

(12) **UK Patent Application** (19) **GB** (11) **2 258 368** (13) **A**
 (43) Date of A publication 03.02.1993

(21) Application No 9220896.6

(22) Date of filing 03.02.1989

Date lodged 05.10.1992

(30) Priority data

(31) 175888

(32) 31.03.1988

(33) US

(62) Derived from Application No 8902396.4 under Section 15(4) of the Patents Act 1977

(71) Applicant

Motorola Inc

(Incorporated in the USA - Delaware)

Corporate Offices, 1303 East Algonquin Road,
 Schaumburg, Illinois 60196, United States of America

(72) Inventors

Kenneth J Zdunek

Jay R Krebs

(51) INT CL⁵

H04Q 7/04

(52) UK CL (Edition L)

H4K KY4D6

(56) Documents cited

None

(58) Field of search

UK CL (Edition K) H4K KYA KYR

INT CL⁵ H04Q 7/00 7/02 7/04

Online databases: WPI, CLAIMS

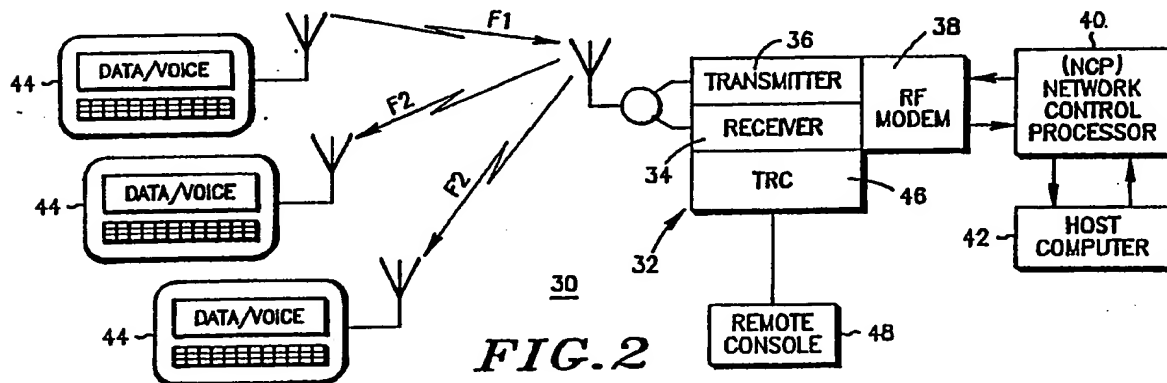
(74) Agent and/or Address for Service

Hugh Christopher Dunlop

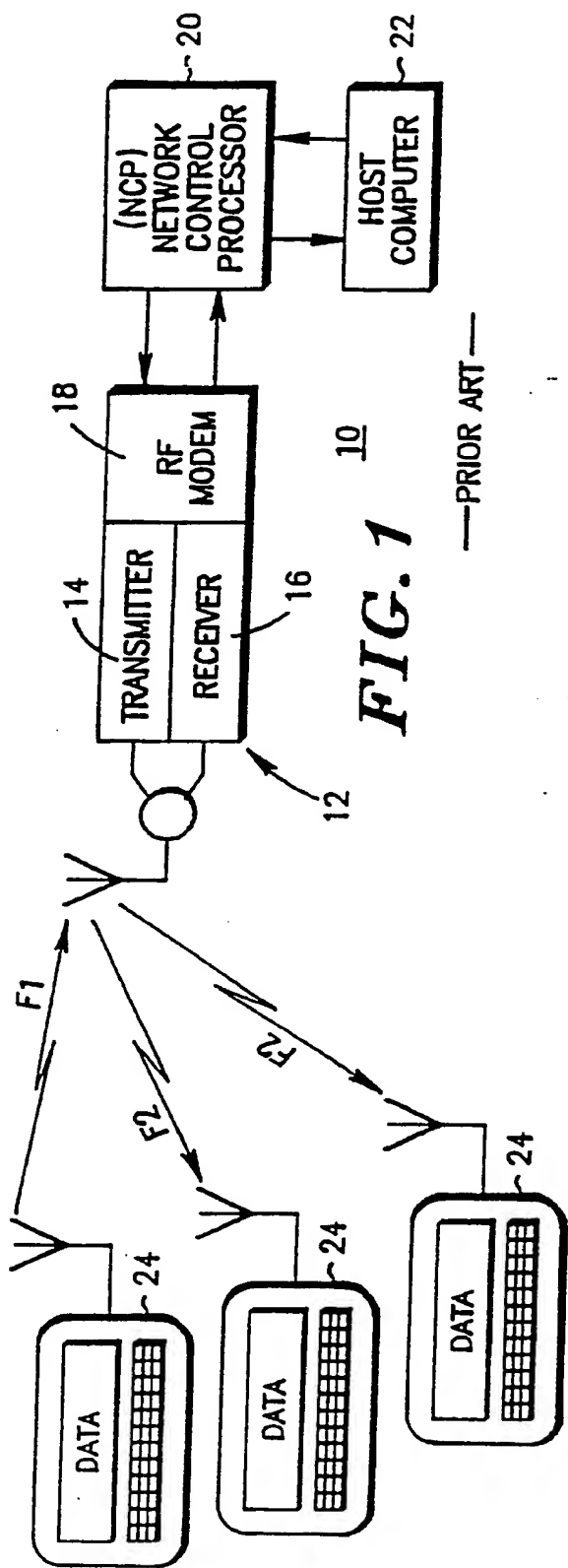
Motorola Limited, European Intellectual Property
 Operations, Jays Close, Viables Industrial Estate,
 Basingstoke, Hampshire, RG22 4PD, United Kingdom

(54) **Data and voice communication system**

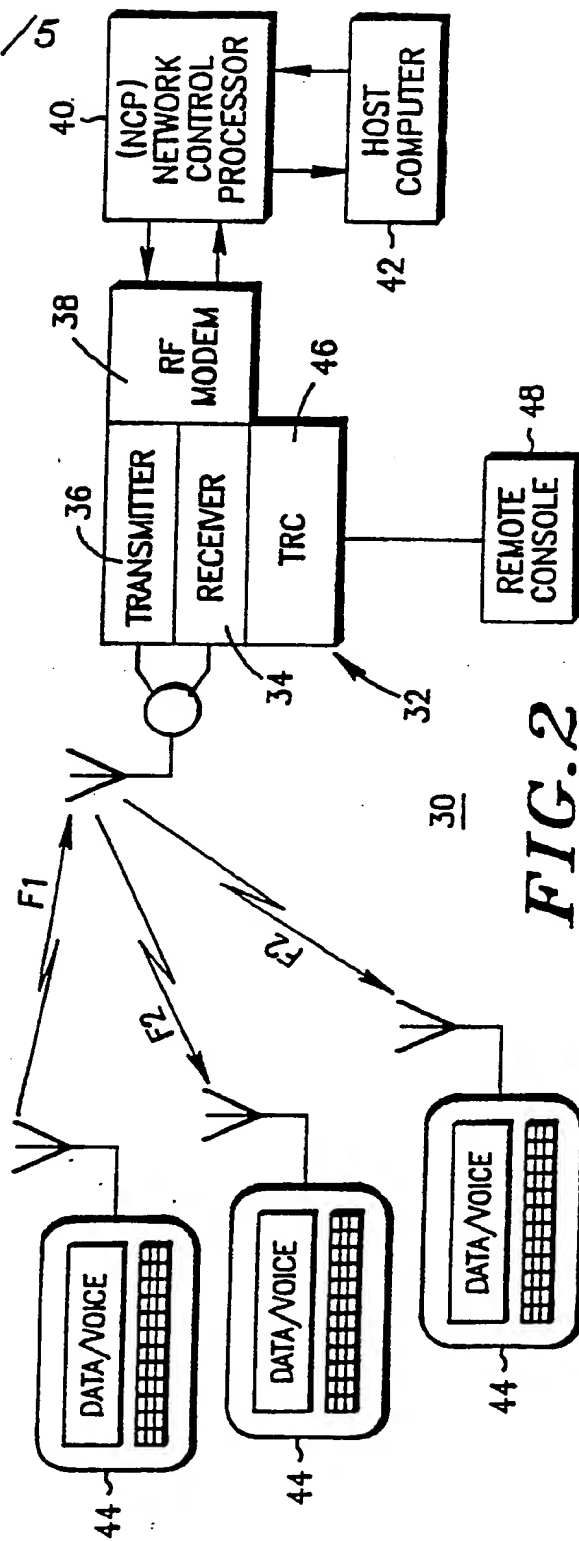
(57) A radio data terminal 44 is capable of operating in voice mode, which mode is requested by transmitting a predetermined data packet and is then initiated on receipt of a subaudible code from a base station. The subaudible code is retransmitted in combination with the voice message. Data traffic may be guaranteed priority at a set, but programmable, level of system capacity so that interference between data and voice traffic is minimized. The system operates on standardized channel access rules in the data mode and when requesting permission to operate in the voice mode, which if granted, effects a specific protocol to condition the system for voice traffic and to manage that traffic. Long and short timers are selectively activated depending upon whether the system is in the voice or data mode to minimize collisions of system radio data terminals requesting channel access.



GB 2 258 368 A



1/5



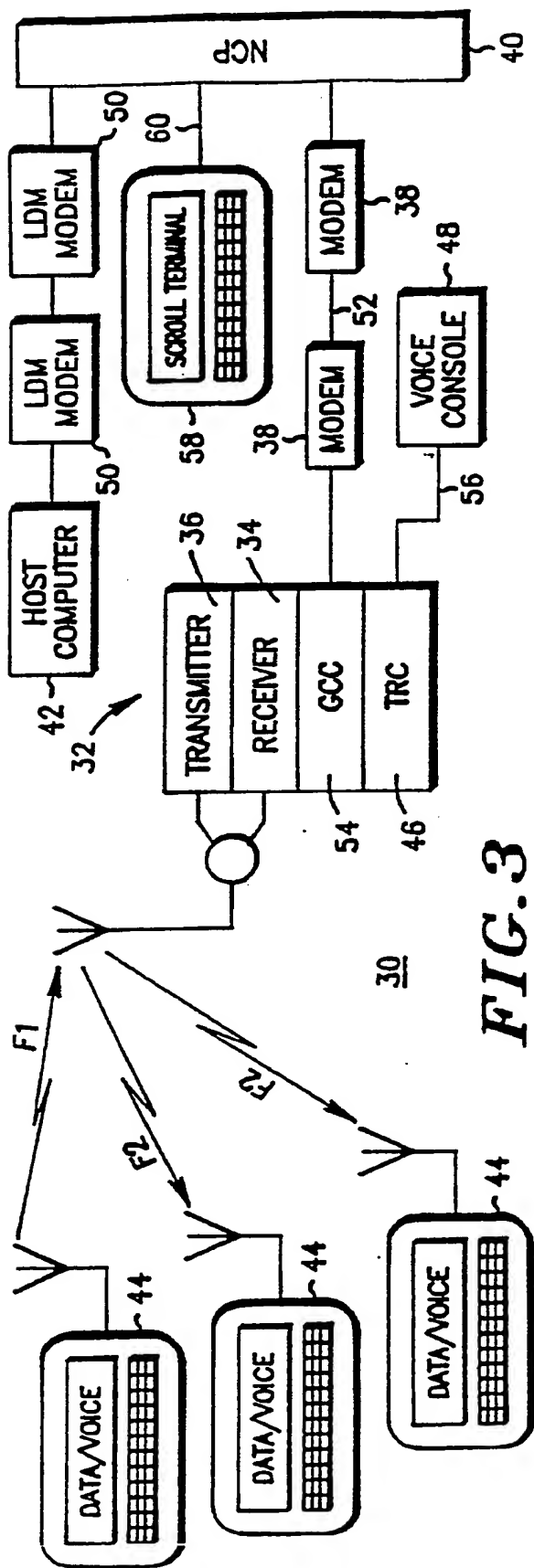


FIG. 3

2/5

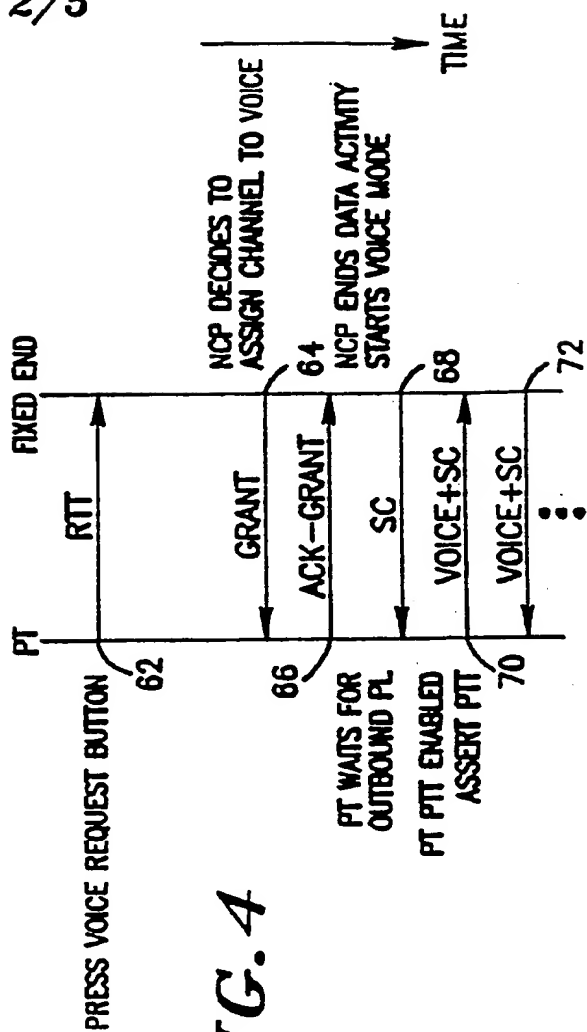


FIG. 4

FIG. 5

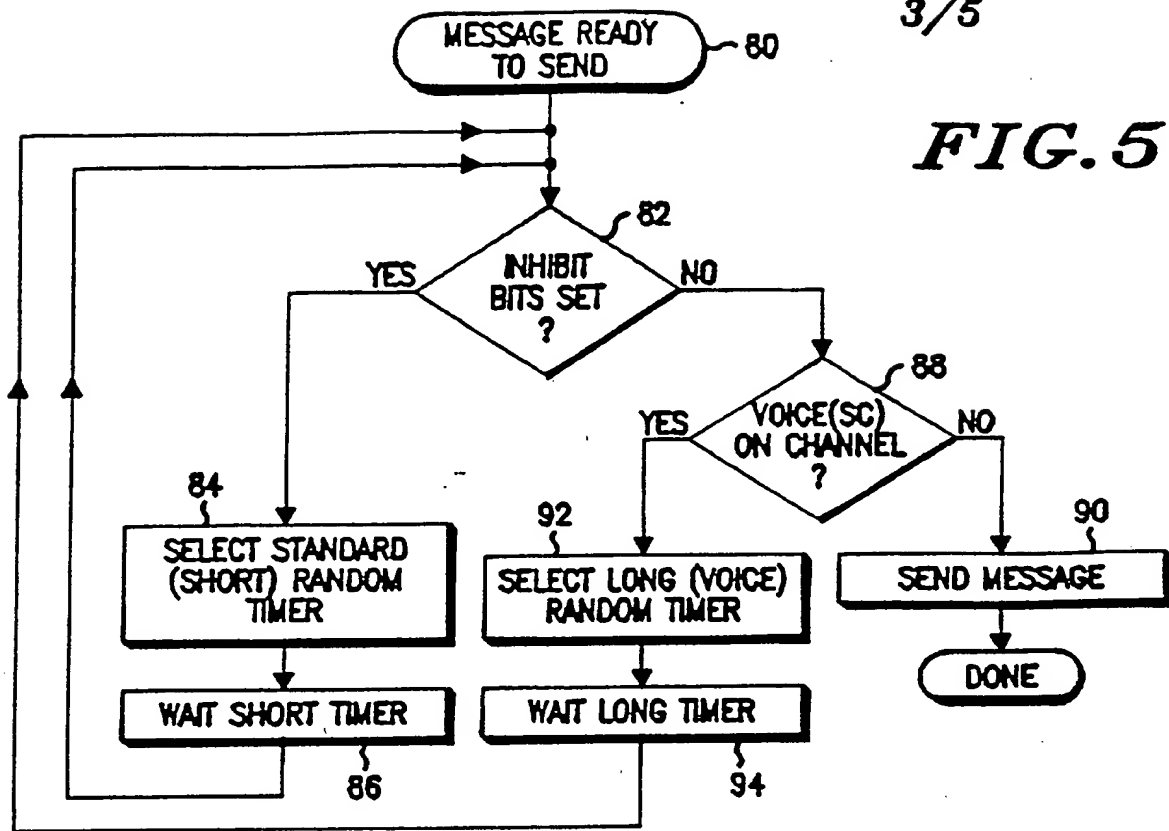
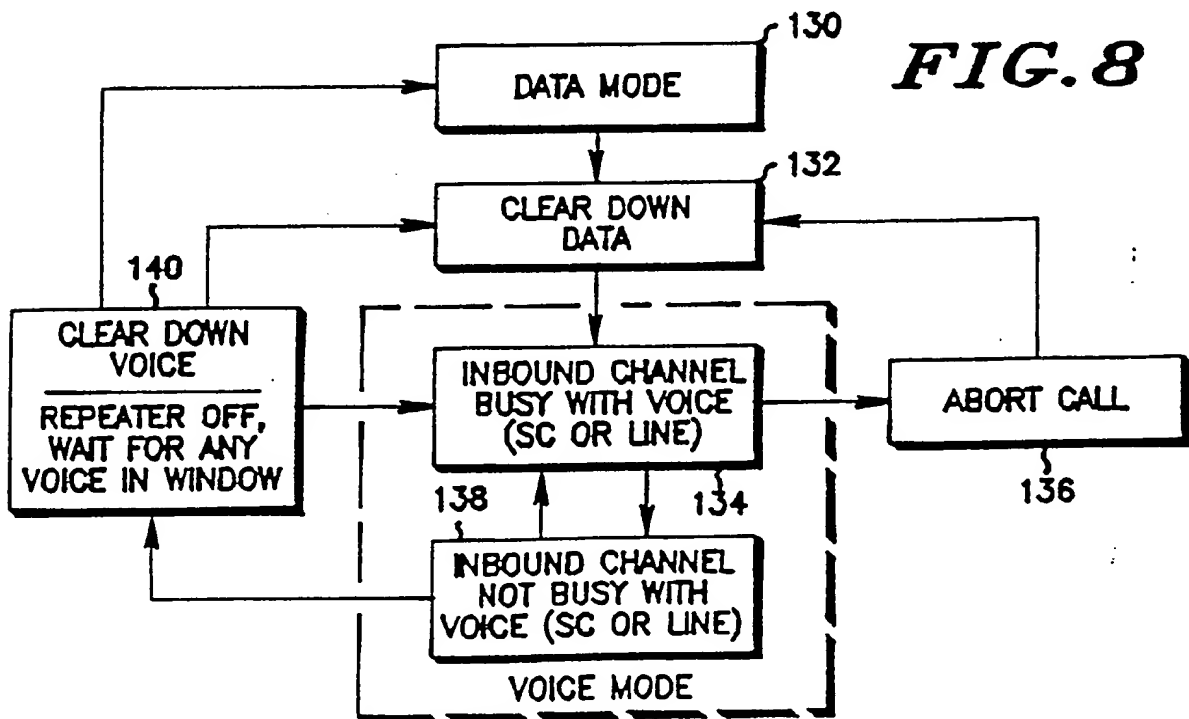


FIG. 8



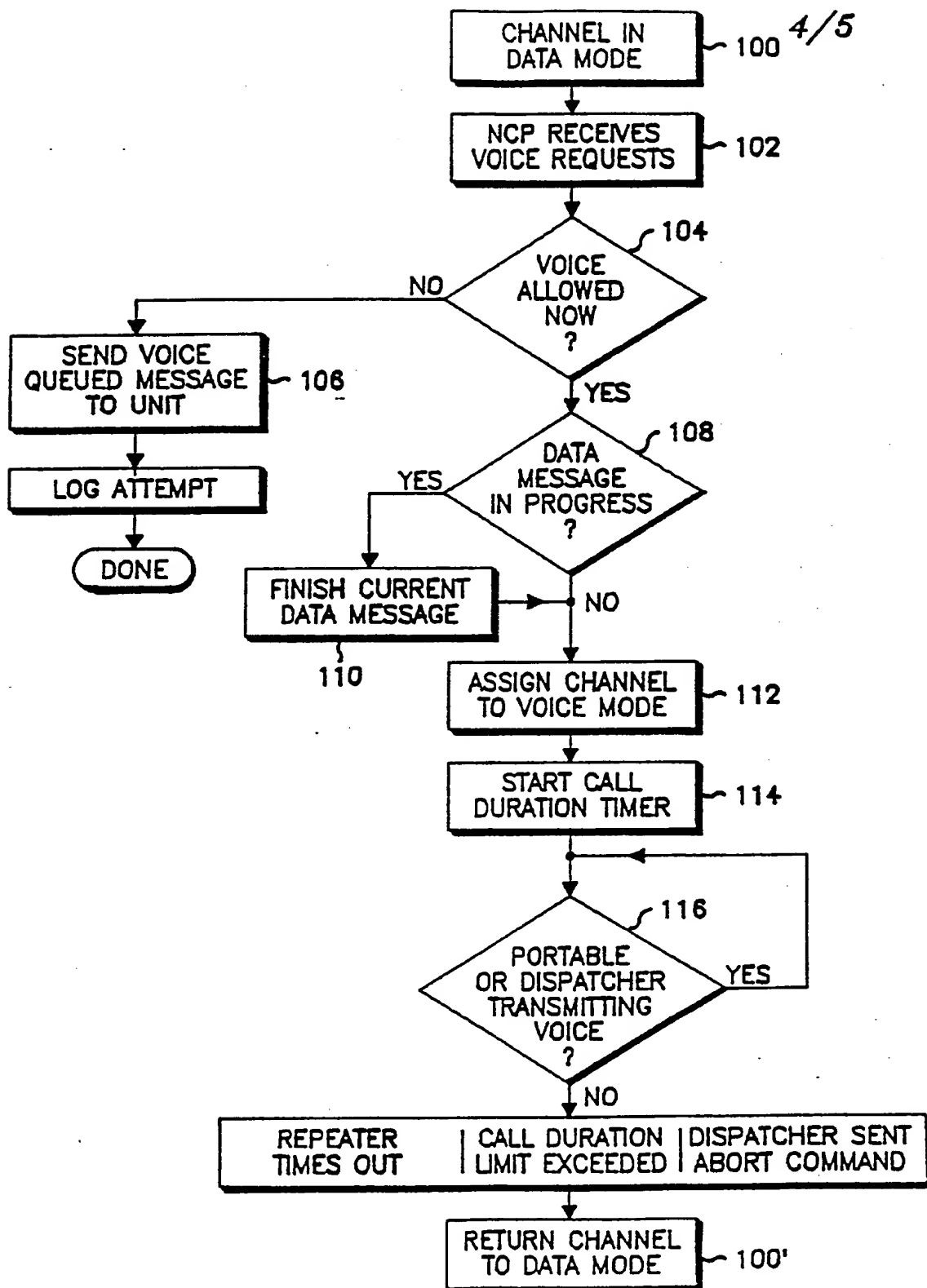
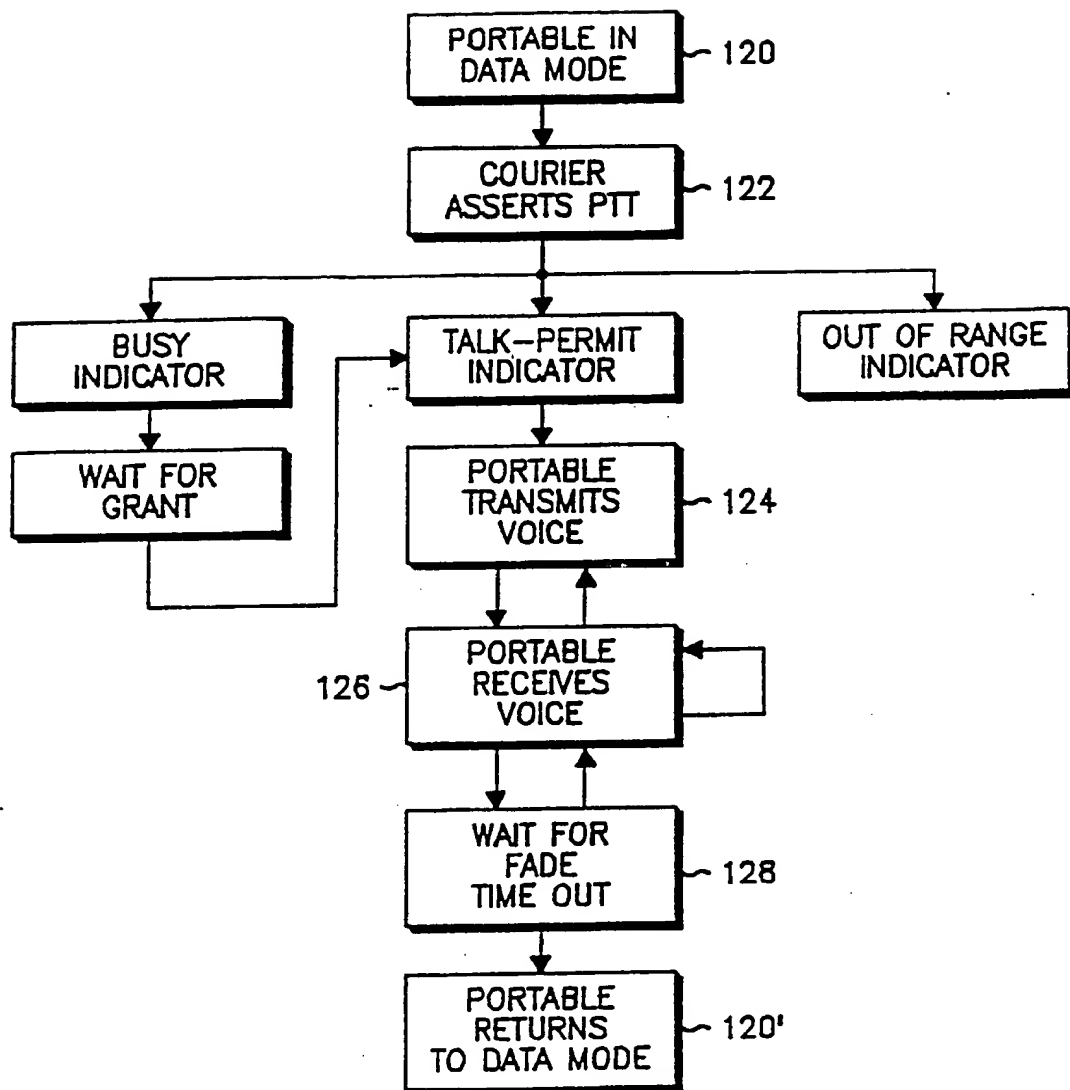


FIG. 6

*FIG. 7*

- / -

SHARED DATA/VOICE COMMUNICATION SYSTEM WITH
PROGRAMMABLE DATA PRIORITY

Background of the Invention

05 This invention relates in general to communication
systems and more particularly to a shared data/voice
communication system wherein the data capacity may be
programmed and maintained as desired and also wherein
interference between voice and data traffic is
10 effectively minimized.

 Communication systems which process data
information are becoming more and more common in the
industry. In point of fact, systems which handle data
15 only are relatively wide spread. Typically such radio
data systems comprise a base station with full duplex
capability, a network control processor (NCP), front end
to a system host computer, and a plurality of portable
radio data terminals operating in half duplex mode.

20 As may be expected, suitable channel access
protocol arrangements are required to minimize terminal
interference and keep the system operating with optimum
efficiency. To this end, a protocol has been developed
which has enjoyed wide spread application, referred to in
25 the literature as "non-persistent busy tone (bit)
multiple access". Basically, the protocol permits
channel contention between radio data terminals. When
one such terminal gains channel access and begins to
transmit data, the base station informs/advises the other
30 system data terminals of such by way of setting "inhibit
bits" at predetermined locations or positions in the
outbound data stream. When a data terminal encounters
these inhibit bits, it waits for a random time before
contending again for access to the channel. In this

way, the system operates with minimum interference and an orderly process is established for all participating data terminals in the system to utilize the available capacity.

05

It should be noted that in such radio data systems, the information being communicated is in fact data. Without more, voice, or analog information, as such is not compatible. There are occasions, however, and increasingly so, where voice communications is very much desired. In some cases, it is absolutely essential. Of course, the user can communicate by voice with a separate mobile or portable unit operating on a completely different channel and/or communication system. As will be readily recognized, such option is quite expensive and, at the same time, inefficient.

There are, of course, systems that accommodate both voice and data on the same system, and even the same channel. However, these are primarily voice systems adapted to also accommodate in limited terms the transmission of data. They are primarily designed to effect channel management with suitable voice protocol arrangements which are not efficient for data management. Moreover, such systems have no means of guaranteeing the system will be used for a minimum level data transmission. Voice communication can and does frequently dominate system usage entirely.

What is needed is the ability to add to an existing efficiently operated data communication system the capability of voice communication as well, without changing the data access protocol being used therein. The addition of the voice communication capability should in no way increase the level of interference on such channel when units are operated in either voice or data

mode. Moreover, there needs to be an effective means and method of guaranteeing that specific but programmable levels of data capacity are being utilized by the system even though it is shared by both the voice and data traffic.

Summary of the Invention

10

In accordance with this invention there is provided a portable radio data terminal intended for operation in a standardized channel access protocol in a data communication system, which terminal is further capable of operating in voice mode, including in combination:

15

means arranged to seek access to the communication channel and request voice mode for the communication system by transmitting a predetermined data packet;

20

means arranged to respond to a command granting access to the channel transmitted to an outbound data stream from a centrally located base station;

25

means arranged to initiate a voice message upon receipt of a subaudible code signal from the base station by retransmitting said code signal in combination with the intended voice message; and

means arranged to terminate said voice mode operation in response to the occurrence of any set, predetermined conditions.

30

In a preferred embodiment of invention, a shared data and voice communication system is provided which has a centrally located base station controlled by a network control processor, as well as a plurality of data terminals with voice capability. The system operates with a predetermined channel access protocol in the data mode and wherein data priority is maintained for a set but programmable level of data usage. The system

35

includes means in each of the data terminals for requesting voice mode for the communication system. And in the network control processor, suitable means are included for granting such requests, inhibiting all other
05 system terminals from sending data messages during the pendency of the requesting terminal operating in the voice mode, as well as means for terminating the voice mode operation at an appropriate time and returning the base station to data mode only operation. The network
10 control processor also includes means for maintaining a set programmable level of data traffic for the communication system by not granting any voice mode requests wherever and whenever the programmed level of data traffic has not been reached as programmed.

15

Brief Description of the Drawings

Figure 1 is a block diagram of a conventional data only communication system which may be considered as known in the art;

20

Figure 2 is a block diagram of a shared voice and data communication system which has been constructed in accordance with the present invention;

25

Figure 3 is a further block diagram in more detail of the communication system of Figure 2 showing a more detailed interconnection between the constituent elements thereof;

30

Figure 4 is a diagramic representation of a request by a portable terminal for a grant by the central network control processor/base station and the associated protocol for operating the communications system in voice mode.

35

Figure 5 is a flow chart of the channel access procedure effected by the system radio data terminals in data voice and data modes;

05 Figure 6 is a flow diagram of the voice call assignment procedure effected by the NCP for granting or not granting a voice request;

10 Figure 7 is a state diagram of a portable radio data terminal operating in the data and voice modes with transition therebetween;

15 Figure 8 is a state diagram of the network control processor and base station operating in the data and voice modes with transition therebetween;

Brief Description of the Preferred Embodiment

Referring now to the drawings, a typical radio data only communication system 10, which may be
20 considered as known in the art, is illustrated in block form in Figure 1. The system has a centrally located base station 12, comprising a transmitter 14 and receiver 16, operating in full duplex mode, which communicates with, by modem 18, and is controlled by, a network
25 control processor (NCP) 20. The NCP in turn communicates with a host computer 22 as indicated. A plurality of portable radio data terminals 24 are also a part of the system 10, operating in half duplex mode.

30 As previously mentioned, a standard channel access protocol has been devised for managing the data communication system 10 of the above type described. The radio data terminals 24 individually seek channel access by transmitting data packets on the inbound channel. The
35 base station/NCP 20 informs/advises the other (and listening) system terminals 24 of this fact by setting

bits located at predetermined positions in the outbound stream. These bits may be referred to as "inhibit bits" and correspond to a conventional busy tone. If another terminal has a data message to be sent, it first monitors
05 the channel and will sense the inhibit bits when set. Upon such occurring, the terminal waits a random time after which it monitors the channel again. This random time may be referred to as the inhibit delay time and is uniformly distributed between zero and some given time T,
10 usually set to substantially correspond to the average inbound data packet length. At some point, perhaps after repeated attempts, the referenced terminal will find or sense no inhibit bits and will be free to transmit its intended data message packet.

15

The system 10 in Figure 1 operates efficiently and effectively, but is capable of data only communication. For the reasons set forth in the background section previously, there is in many instances a need for voice
20 capability as well. Note that it is a need for voice capability in essentially a system designed for processing data and not vice versa. It is not a system intended for voice communications only at the inception and, subsequently, where the capability of processing
25 data is added.

The data system of the present invention, which permits limited voice communications, is set forth diagrammatically in Figure 2, which system has been
30 constructed in accordance with the precepts of the present invention. In broad terms, the system 30 of Figure 2 is much like that of Figure 1 in that a centrally located base station 32 is provided comprised of a receiver 34 and transmitter 36. Modem 38 permits
35 communication with the NCP 40 which in turn communicates with a host computer 42. A plurality of portable/mobile

radio data terminals 44 are also in integral part of the system. In this case, radio data terminals 44 are also capable of voice transmission and have a microphone (not shown) with a push-to-talk switch. Further, as a part of the arrangement for effecting voice capability, the tone remote console (TRC) 46 and remote console 48 is provided.

A more detailed representation of the system 30 is shown in Figure 3. In addition to the previously referenced component parts, system 30 includes limited distance modems (LDM) 50 permitting communications between the NCP 40 and host computer 42 on a full duplex basis and are of the type referenced as HDLC ABM modems operating with the capability of 9.6 kilobits. A general communications controller (GCC) 54 is located at the base site intercoupled to the NCP through associated modems. Modem 38 of Figure 2 is shown as a pair of modems in Figure 3, which are also of the HDLC ABM type with 9.6 kilobits capability. Wire lines indicated at 52 constitute a four-wire TELCO circuit to permit full duplex communications between base station 32 and NCP 40. A two-wire, voice grade, line 56 connects the TRC 46 with the remote console 48. Finally, a scroll mode terminal 58 is provided which is coupled by an RS-232 link 60 to the NCP 40. Data information is passed between the NCP 40 and the base station 32 via modems 38 and wire line 52 while voice communication is processed by the dispatcher at console 48 over wire line 52 to the base station 32. Terminal 58 permits the dispatcher to keep track of the data or voice mode the system is operating on as well as the identity of the user utilizing the system at any particular moment in time.

It will be recalled that system 32 is primarily designed to process data. Further, it operates with a

specific channel access protocol as previously described in managing the data traffic over the system. Even though system 32 has been given the capability of voice transmission in accordance with the present invention, it is to retain such channel access protocol for handling the data but make provision for permitting and managing voice traffic as well. Accordingly, in the first instance, the portable radio data terminal 44 must request authorization to operate in the voice mode by first sending a request to talk (RTT) in the form of a data packet essentially in the same way as it does for data. Upon receipt at the base station 32, a decision by the NCP 40 must be made whether to grant such request. If in the affirmative, a suitable protocol arrangement must be initiated to grant the request and manage the ensuing voice communication while keeping other radio data terminals from attempting to communicate during the pendency of such voice message.

This is depicted graphically in Figure 4 showing the necessary steps to initiate voice communication by a particular radio data terminal 44 and the ensuing protocol to establish the same and manage the system. The radio data terminal 44 initiates the request by pushing a voice request button on the terminal at step 62 which effects transmission of the appropriate data packet interpreted as a request to talk (RTT). Upon receipt at the base station and routed to the NCP a decision is made by it whether or not to assign the system channel to voice. The factors for making this decision will be discussed subsequently. If the decision is in the affirmative, the grant is made at step 64. Upon receipt by the portable radio data terminal 44, it acknowledges back the grant at step 66.

35

The data terminal then waits for further action by the NCP 40. The NCP ends all data activity on the channel and then initiates voice mode by first transmitting a supervisory, sub-audible code signal (SC) at step 68. The sub-audible code may be either a tone or a digital signal as desired. Upon receipt thereof by the radio data terminal 44, voice communication may commence and is effected upon the push-to-talk switch (not shown) being activated at step 70. This causes the voice message to be transmitted along with the subaudible code signal. If the dispatcher wishes to send a voice message back to the radio data terminals, the same is likewise transmitted along with the subaudible code signal as indicated at step 72.

15

The voice activity continues until completed or is terminated by predetermined conditions. In general, voice activity may be terminated by any of the following: (a) overall call length time limit being exceeded; (b) voice inactivity time out being reached; (c) being terminated at the option of the dispatcher at his or her discretion; or (d) by being preempted by an emergency call.

25

In this manner, the overall system 30 operates with either data or voice, but continues to function effectively using the standard data channel access rules. The system radio data terminals 44 may not transmit voice until a request is made to do so using a data message indicating such request. When, and only when, such request is granted, may the requesting terminal transmit voice, and even here, only after the NCP converts the system to voice mode operation and indicates such by transmitting a supervisory, subaudible code signal. The other radio data terminals in the system are prevented from accessing the channel during the pendency of either

35

a data or voice message. While data is on the system channel the inhibit bits are set in the outbound data stream as previously described and will cause another radio data terminal when it monitors the channel to wait
05 a random time before it attempts access again. Likewise, if a data terminal, when monitoring a channel, encounters voice, such terminal will again wait a random time before attempting to access again. It should be pointed out, however, since voice activity inherently results in a
10 longer use of the channel, that the random times waited by other radio terminals attempting access must necessarily be longer when the system is operating with voice communication. In other words, the inhibit times should be longer for voice operation than for data. This
15 is depicted in the flow chart as shown in Figure 5.

As shown therein, when a message is ready to be sent by a radio data terminal 44 at step 80, the channel is monitored to determine if the inhibit bits are set at
20 step 82. If yes, indicating the channel is in use for data operation, the standard (short) random inhibit timer is activated at step 84, such timer being in the range of 0 to 250 milliseconds. The terminal seeking access waits a random time within this range at step 86 and attempts
25 channel access again at step 80. Conversely, if inhibit bits are not detected as being set (step 82) the terminal determines if voice is on the channel at step 88 and, if not, sends its message in the usual manner. If the voice is detected, the terminal sets the long (voice) random
30 timer at step 92, in the range of 0 to 10 seconds, and waits a random time within that range at step 94 before again attempting channel access at step 80.

In this way, the number of messages that may
35 otherwise be lost due to repeated collisions for channel access may be significantly reduced when the system is

operating in the voice mode, because the "long" inhibit timer is invoked rather than the standard or short term timer utilized when in data mode since the data messages are normally much shorter in duration than voice.

05

As mentioned previously, one of the advantages of the system of the present invention is that it maintains a user specified maximum percent utilization of voice on the channel. This guarantees the channel will be reserved for a given level of data activity. As an example, if the maximum amount of voice traffic is chosen not to exceed 15 percent, the system will be conditioned to guarantee that 85 percent of the available time will be devoted to data traffic. Of course, this ratio can be changed as needed or simply preferred.

15

This selection of setting the ratio of data to voice utilization is an important aspect in the decision by the NCP 40 in determining whether or not to immediately grant authorization or voice communication when a request is received from a system radio data terminal 44, or, if not, to queue the request for later grant on a first in, first out basis when the request may be honored.

25

This decision-making process and procedure is depicted in the flow chart as shown in Figure 6. The algorithm as therein contained determines the voice call assignment procedure as well as maintaining the specified ratio of data to voice traffic. When a period of voice activity ends, the NCP 40 temporarily stores the duration of the voice message. It then computes a "data interval" based on this voice message duration and the desired or specified maximum utilization of the channel for voice. This data interval may be determined as follows:

35

Data Interval = $\frac{(1 - \text{Max. Voice Util.})}{\text{Max. Voice Util.}} \times \text{Voice Message Duration.}$

Any voice RTT received during this data interval (steps 102, 104) is then queued, step 106, for later
05 processing on a FIFO basis when voice is permitted.
There is, of course, a minimum length to the "data interval" of several seconds to handle any data activity which may be pending as a result of voice being on the channel. This minimum data interval can be a fixed
10 length based on estimates of the data traffic on the system, or it may be dynamically adjusted according to the NCP estimate of current "penned-up" data traffic. This may, for example, be derived from the NCP out bound message queues.

15

It will be appreciated that, if the situation is such that voice may be allowed at step 104, the NCP 40 then checks to see if any data message is in progress at step 108 and, if so, will delay until the current data
20 message is finished at step 110. If there is no data message in process, the assignment of the channel to voice mode may then be made at 112 as described previously. The start call duration timer may be then initiated at step 114 if there is no portable or
25 dispatcher transmitting voice at step 116. The voice communication continues until such time as the repeater times out or the call duration limit has been exceeded or the dispatcher sends an abort command to terminate. Following the cessation of voice communication, the
30 channel is then returned to data mode again at step 100'.

The portable function on voice mode is depicted in the state diagram of Figure 7. As shown, the portable while in the data mode at step 120 may assert the PTT to
35 request permission for voice communication at step 122

and either receives from the centrally located base station 32 a busy indication or a grant from the central base station for permission to use voice communication.

05 If a busy indication is first received, it will be understood that the data terminal will await an appropriate time after which the NCP 40 may determine that the grant may be made. If so, it will be made in the manner previously discribed. If the signal from the
10 NCP 40 cannot be received and processed, an out of range indication will be initiated. As previously indicated, the portable waits for a subaudible coding signal in the outbound message stream to authorize transmission by the portable of voice communication as indicated at step 124.
15 The portable may receive voice by releasing the PTT and monitoring the channel at step 126 and may reassert the PTT to further transmit voice communication by reverting back to step 124. If, however, the portable, in
20 monitoring the channel for voice transmission, fails to receive the same within a specified period of time as indicated at step 128, it simply returns to the data mode as indicated at step 120'.

 A state transition function diagram for the fixed
25 end is indicated in Figure 8. Assuming that the channel for system 30 is in the data mode at step 130, it will remain in such mode until a request to voice communicate is received from one of the system radio data terminals
30 44. Upon deciding that such voice may be allowed at that point, the NCP 40 sends out the grant and upon receiving an acknowledgment back initiates the clear down data function at step 132 where it may complete any current
35 outbound message, cause the GCC 46 to be put in the "inhibit-on" mode as well as sending any queued acknowledgments and ensure that the subaudible coding is activated and transmitted on the outbound stream. It

then waits for a detect of the subaudible coding or line PTT to indicate that the inbound channel is busy with voice as indicated at step 134. Thereafter the call may be aborted by the dispatcher or by the call duration
05 limit being exceeded, as indicated at step 136, which then causes the repeater to shut down and it will revert to the data mode at step 130 when the portable data terminal is no longer on the channel. Alternatively, if after detecting voice on the channel at step 134, and the
10 call is not specifically aborted, the NCP 40 will monitor the inbound channel, and if subaudible coding or no line PTT is detected, the inbound channel will be determined as not busy and may either, after a repeater drop out time has expired, revert to the clear down voice mode at
15 step 138, or, if the subaudible coding is received before that time, revert back to step 134 where the inbound channel is determined as active and voice is on the channel. Assuming that step 138 has been reached, the repeater will then be turned off and the system will wait
20 for any voice in a predetermined window after which it is returned to the data mode at step 130.

Accordingly, a shared data/voice communication system has been described wherein interference between
25 data and voice mode operation is effectively minimized while at the same time guaranteeing that a set level of system capacity is reserved for data only traffic. The system operates without change on standard data channel access rules or protocol at all times except when a voice
30 transmission has been authorized and is in fact taking place, wherein the specialized voice management protocol takes effect. Long and short delay timers are included in the portable radio data terminals regarding random times the terminals are to wait, depending on whether the
35 system is in voice or data mode, before attempting channel access and thereby minimize message interference

as a result of excessive collisions when attempting such
access. In this way, any system radio data terminal may
request, and when granted, communicate a voice message to
the central base station. In like manner, portable-to-
05 portable selective calls may be initiated as well as
portable initiated group calls, dispatcher initiated
selective calls, dispatcher initiated group calls,
portable initiated emergency calls when the channel is in
data mode, and portable initiated emergency calls when
10 the channel is in voice mode.

15

20

25

30

35

CLAIMS

- 5 1. A portable radio data terminal intended for operation in a standardized channel access protocol in a data communication system, which terminal is further capable of operating in voice mode, including in combination:
- means arranged to seek access to the communication
10 channel and request voice mode for the communication system by transmitting a predetermined data packet;
- means arranged to respond to a command granting access to the channel transmitted on an outbound data stream from a centrally located base station;
- 15 means arranged to initiate a voice message upon receipt of a subaudible code signal from the base station by retransmitting said code signal in combination with the intended voice message; and
- means arranged to terminate said voice mode operation in
20 response to the occurrence of any set, predetermined conditions.
2. A portable radio data terminal in accordance with claim 1 wherein said means arranged to seek access to the communication channel includes means arranged to first determine if inhibit bits
25 are set in the outbound stream on the communication channel, and if so, to initiate a short random inhibit timer, and if not, to either initiate a long random inhibit timer, if voice is detected on the communication channel, or send its intended message if voice is not detected on the communication channel.
- 30 3. A portable radio data terminal in accordance with claim 2 wherein the short random timer is in the range of 0-250 milliseconds.
- 35 4. A portable radio data terminal in accordance with claim 2 or 3 wherein the long random timer is in the range of 0-10 seconds.

5. A portable radio data terminal in accordance with any one of claims 1 to 4 wherein said means arranged to terminate voice mode operation includes means arranged to operate in response to either of the call duration limit being exceeded, the base station timing out or the dispatcher transmitting an abort command.

- 10 -

Patents Act 1977
Examiner's report to the Comptroller under
Section 17 (The Search Report)

Application number
GB 9200896.6

Relevant Technical fields

- (i) UK Cl (Edition **K**) **H4K (KYA, KYR)**
- (ii) Int Cl (Edition **5**) **H04Q 7/00, 7/02, 7/04**

Search Examiner

G N CHAPMAN

Databases (see over)

- (i) UK Patent Office
- (ii) **ONLINE DATABASES: WPI, CLAIMS**

Date of Search

27 OCTOBER 1992

Documents considered relevant following a search in respect of claims

1 TO 5

Category (see over)	Identity of document and relevant passages	Relevant to claim(s)
	NONE	

Category	Identity of document and relevant passages	Relevance to claim(s)

Categories of documents

X: Document indicating lack of novelty or of inventive step.

Y: Document indicating lack of inventive step if combined with one or more other documents of the same category.

A: Document indicating technological background and/or state of the art.

P: Document published on or after the declared priority date but before the filing date of the present application.

E: Patent document published on or after, but with priority date earlier than, the filing date of the present application.

&: Member of the same patent family, corresponding document.

Databases: The UK Patent Office database comprises classified collections of GB, EP, WO and US patent specifications as outlined periodically in the Official Journal (Patents). The on-line databases considered for search are also listed periodically in the Official Journal (Patents).